Advanced Network Technologies

Week 2:

Network performance

Network application

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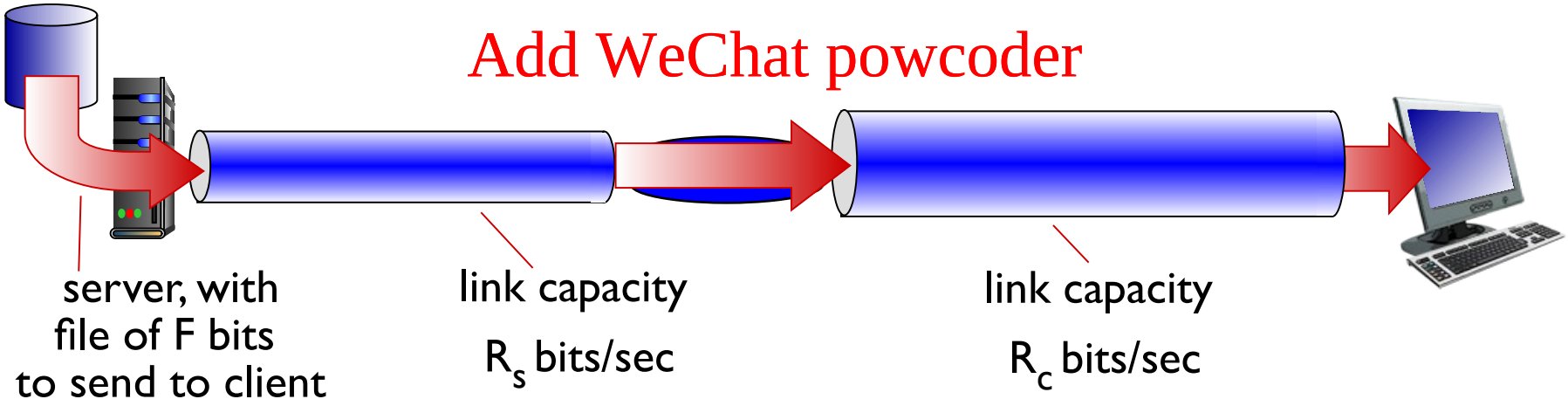
Network Performance: Throughput

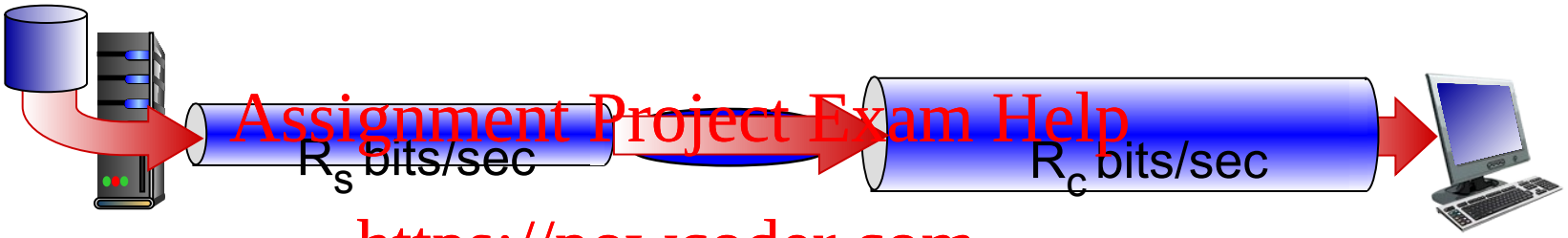
Throughput

- *throughput:* rate (bits/time unit) at which bits transferred between sender/receiver
  - *instantaneous:* rate at given point in time
  - *average:* rate over longer period of time



Throughput (cont'd)

› $R_s < R_c$ What is average end-end throughput?



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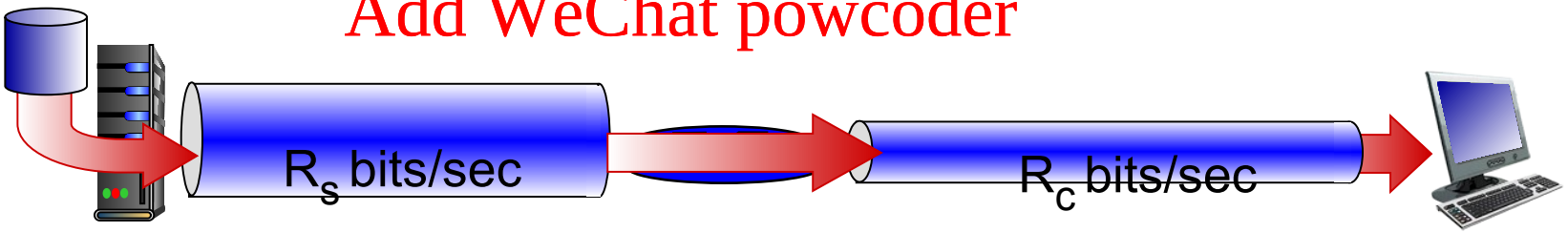


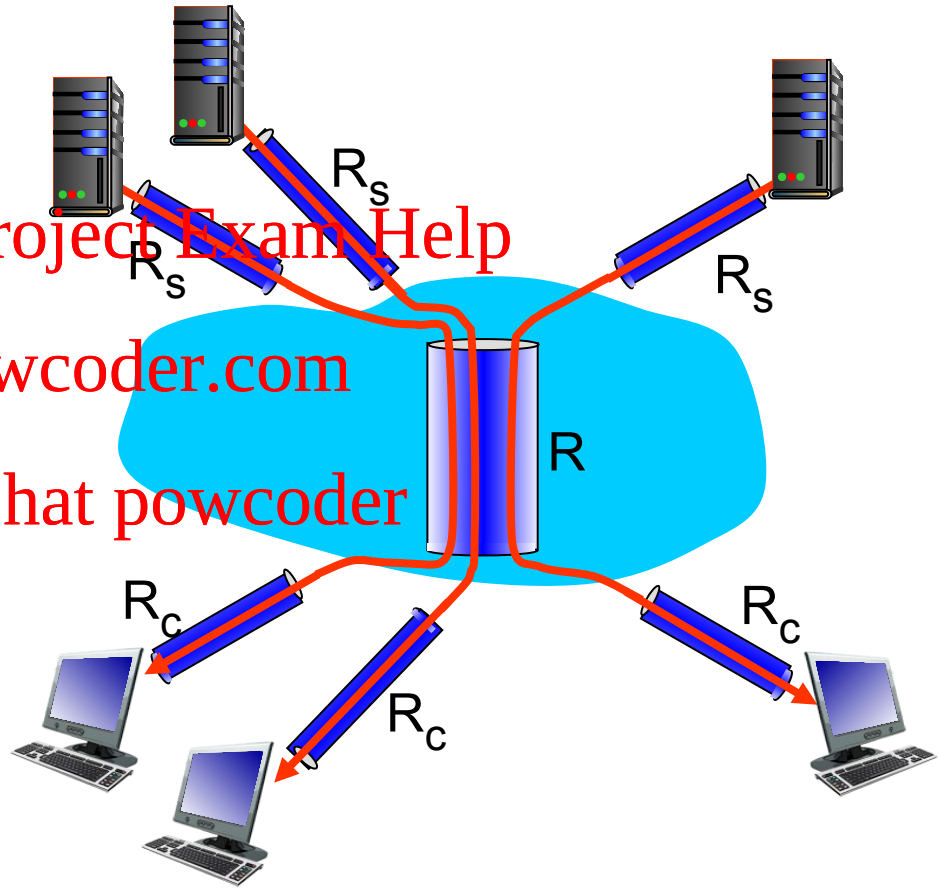
bottleneck link

link on end-end path that constrains end-end throughput

Throughput (cont'd)

Internet Scenario

- per-connection end-end throughput: min(R_c,R_s,R/10)

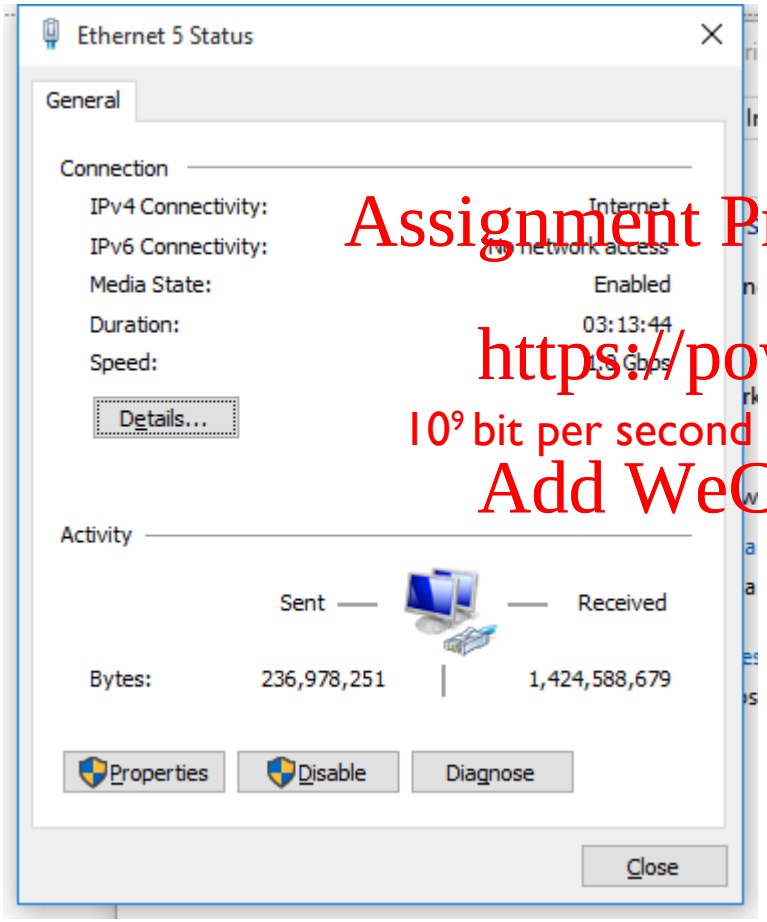


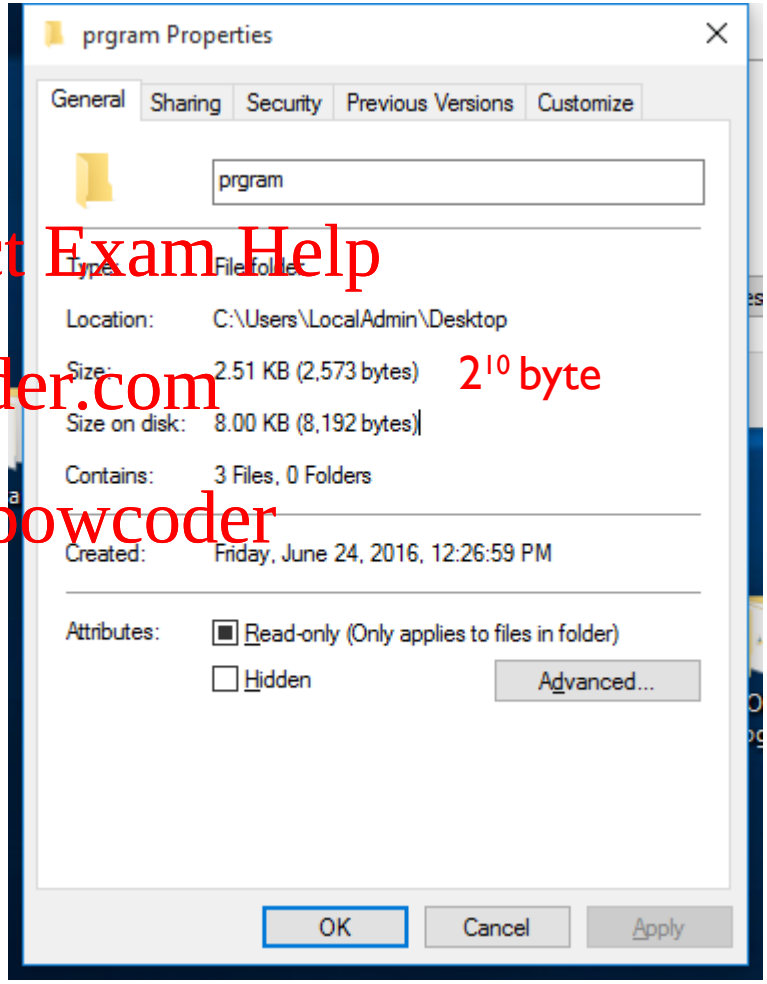
10 connections (fairly) share backbone bottleneck link R bits/sec

Bit and byte

- bit: basic unit. "b"
- byte: 8 bits. "B"
- bps: bit per second
- Network/Telecom:
  - Kb/Mb/Gb: $10^3, 10^6, 10^9$ bit
  - Kbps/Mbps/Gbps: $10^3, 10^6, 10^9$ bit per second
  - By default in this course
- File system:
  - KB/MB/GB: $2^{10}, 2^{20}, 2^{30}$ byte ($1024, 1024^2, 1024^3$ byte)

Bit and byte





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Network Performance: Fairness

Network Fairness and Bandwidth Allocation

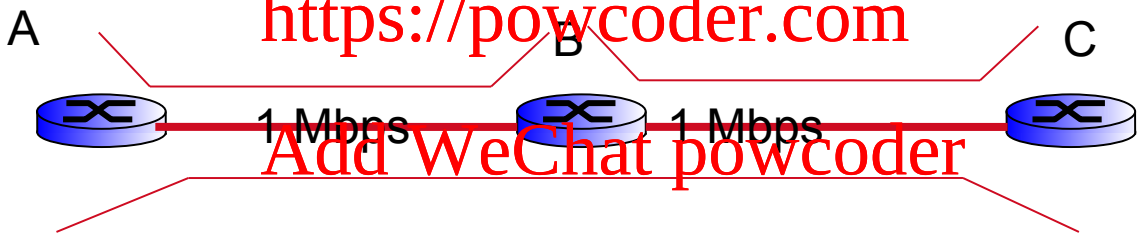
In reality: two considerations

- Efficiency
- Fairness

- However, they are contradicting!

Network Fairness, Bandwidth allocation

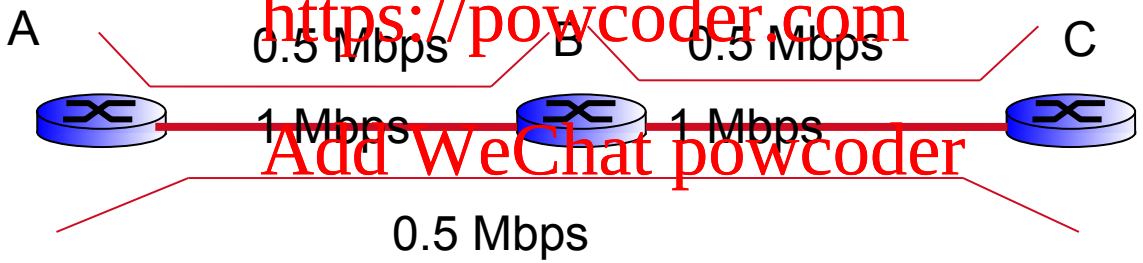
Three flows: A-B, B-C, A-C



Q: How can we allocate the link bandwidths to the three flows?

Network Fairness, Bandwidth allocation

Three flows: A-B, B-C, A-C

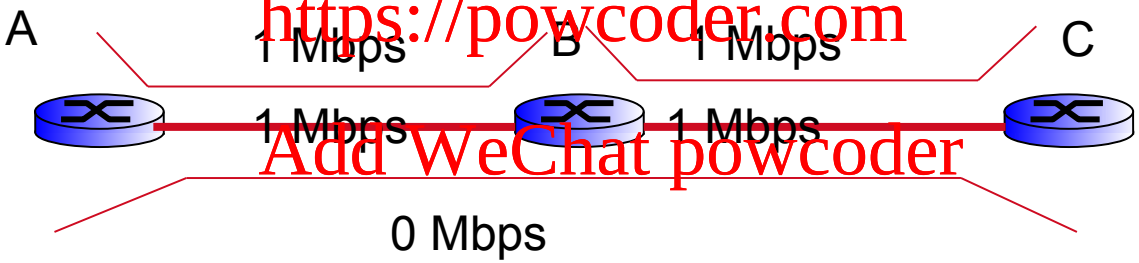


Very fair!

However: Network throughput, only 1.5Mbps

Network Fairness, Bandwidth allocation

Three flows: A-B, B-C, A-C

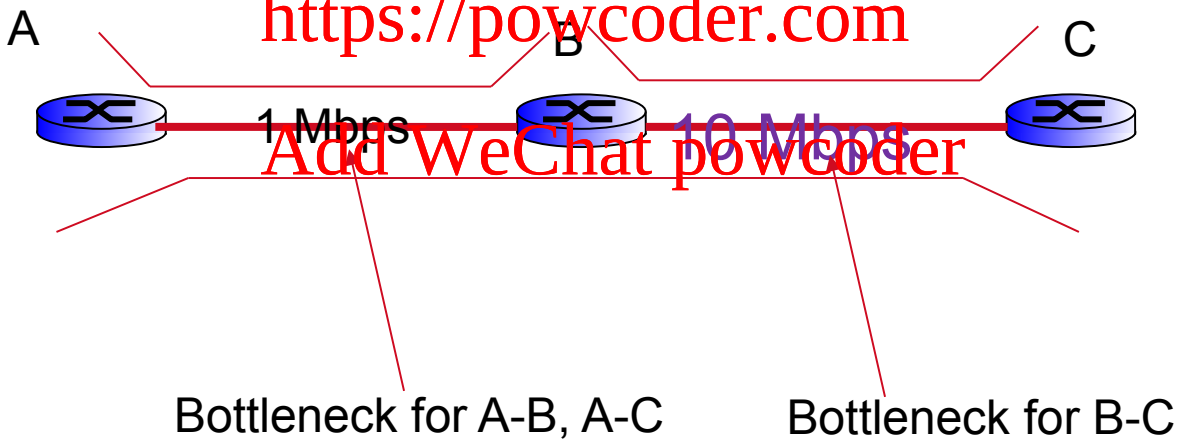


Very unfair!

However: Network throughput, 2Mbps

Fairness

Bottleneck for a flow: The link that limits the data rate of the flow



Max-min Fairness

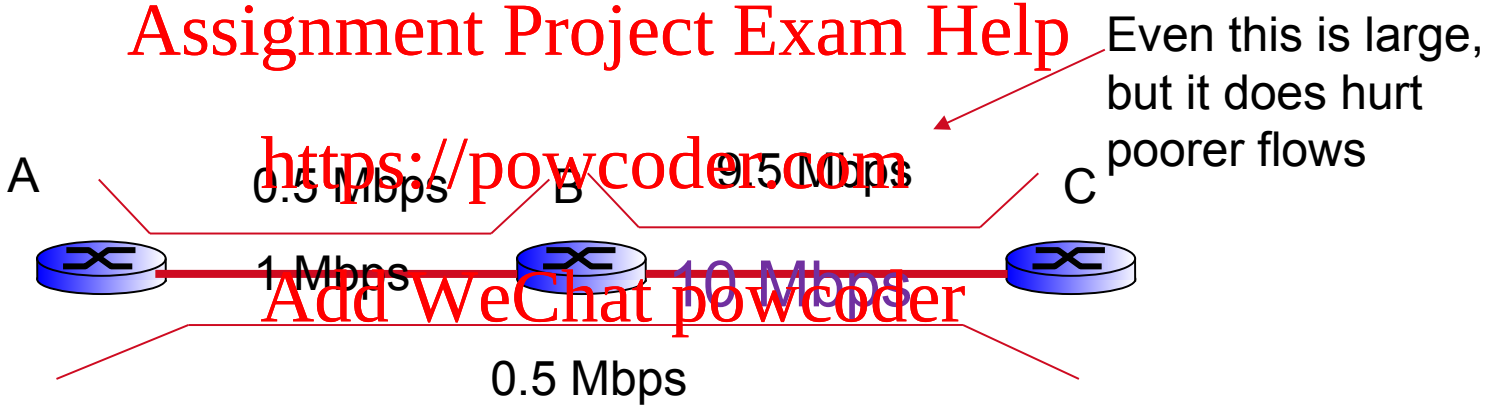
- Maximize the minimum
- Try to increase the "poorest" as much as possible
  - A richer flow can be sacrificed.
- Try to increase the second "poorest" as much as possible
  - A richer flow can be sacrificed.
  - A poorer flow cannot be sacrificed.
- Try to increase the third "poorest" as much as possible
- …

Max-min Fairness

- Max-min Fairness criteria: if we want to improve one flow, we can only achieve this by sacrificing a poorer or equal flow.

Max-min Fairness

Bottleneck for a flow: The link limits its data rate



Bottleneck approach

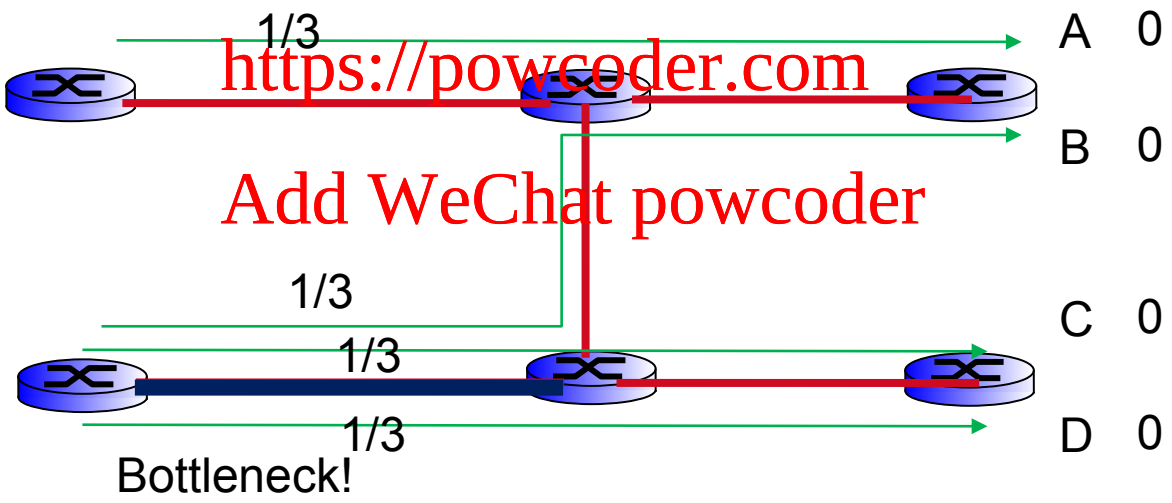
Bottleneck approach

- 1 Start with all zero flows, potential flow set = {all flows}
- 2 Slowly increase flows in the potential flow set until there is a (new) link saturated
  - "Pouring water in the network"
- 3 Hold fix the flows that are bottlenecked, remove them from the potential flow set
- 4 If potential flow set is not empty, go to step 2 (still has potential to increase)

Bottleneck approach

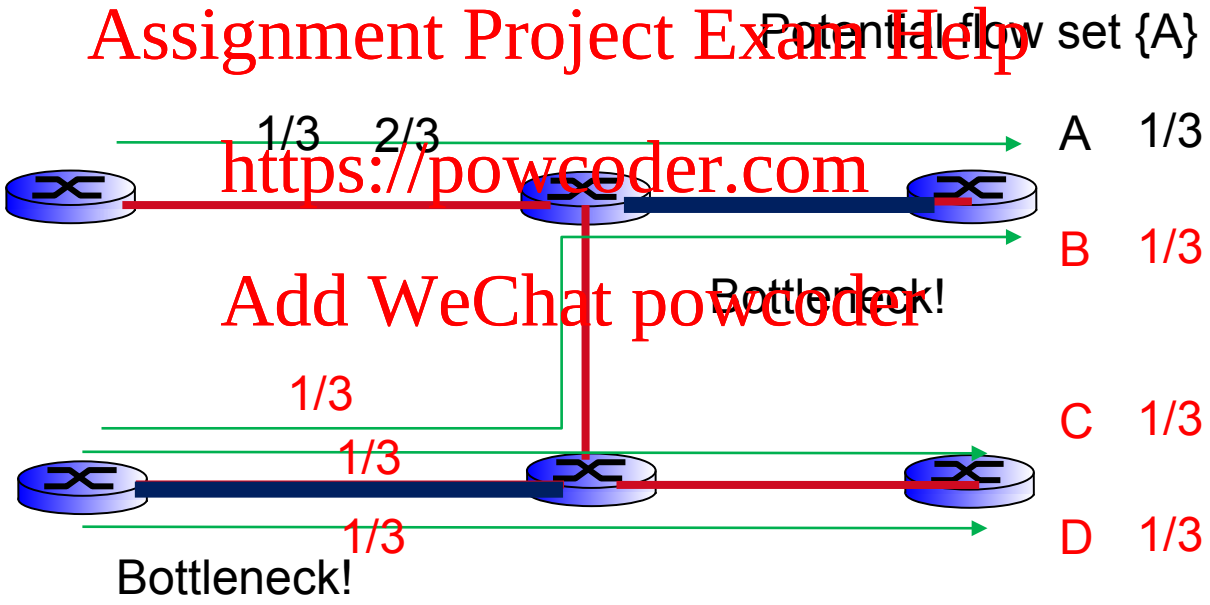
Each link between two routes with capacity 1

Potential flow set {A, B, C, D}



Bottleneck approach

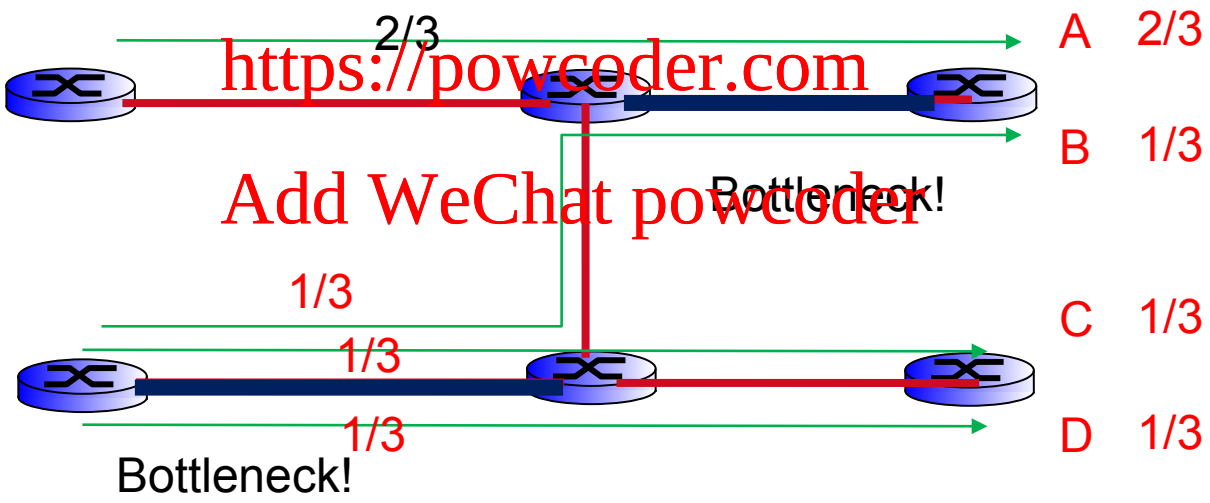
Each link between two routes with capacity 1



Bottleneck approach

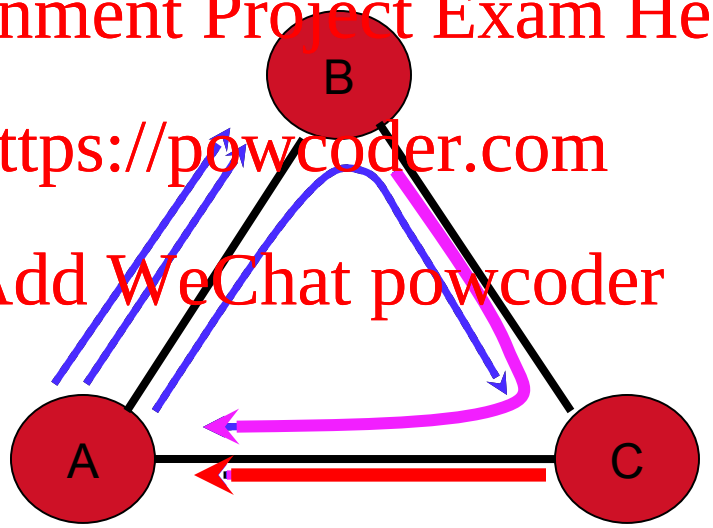
Each link between two routes with capacity 1

Potential flow set {}



Can you solve the following problem?

link rate: AB=BC=1, CA=2

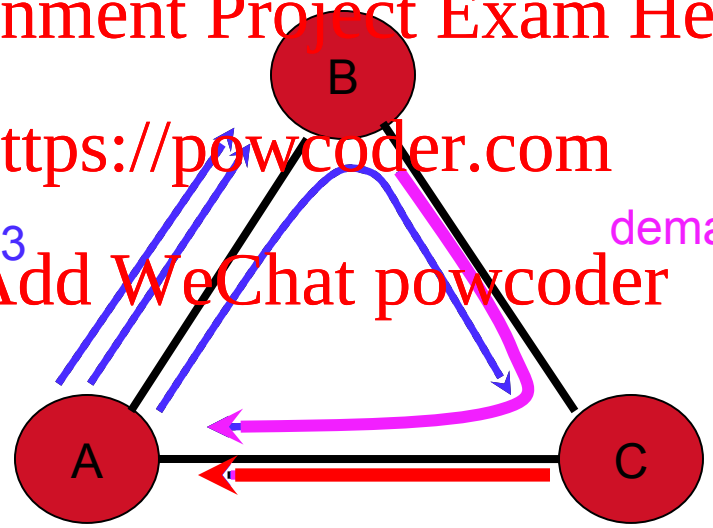


Can you solve the following problem?

link rate: AB=BC=1, CA=2

demand 1,2,3 =1/3

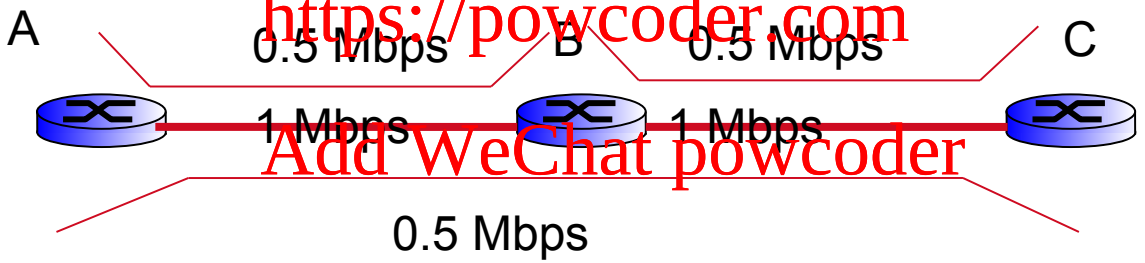
demand 4 =2/3



demand 5=4/3

More comments

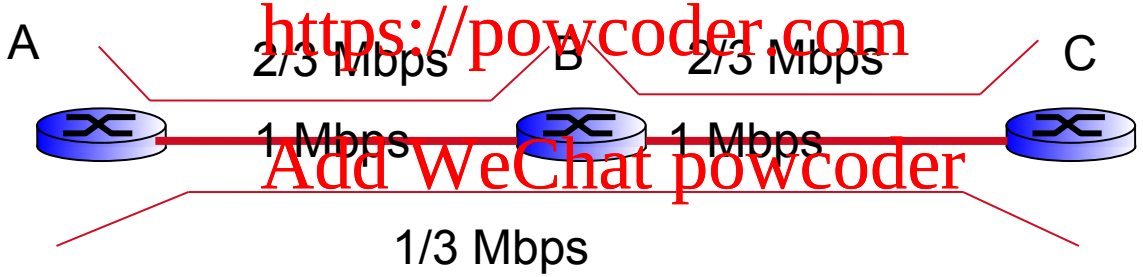
More comment: Max-min fairness is too fair!



You are using two links. How can we get a same share?

Another form of fairness

proportional fairness



Longer routes are penalized

The Application Layer

Some network applications

- e-mail
- web
- text messaging
- remote login
- P2P file sharing
- multi-user network games
- streaming stored video (YouTube, Netflix)
- voice over IP (e.g., Skype)
- real-time video conferencing
- social networking
- search
- …
- …

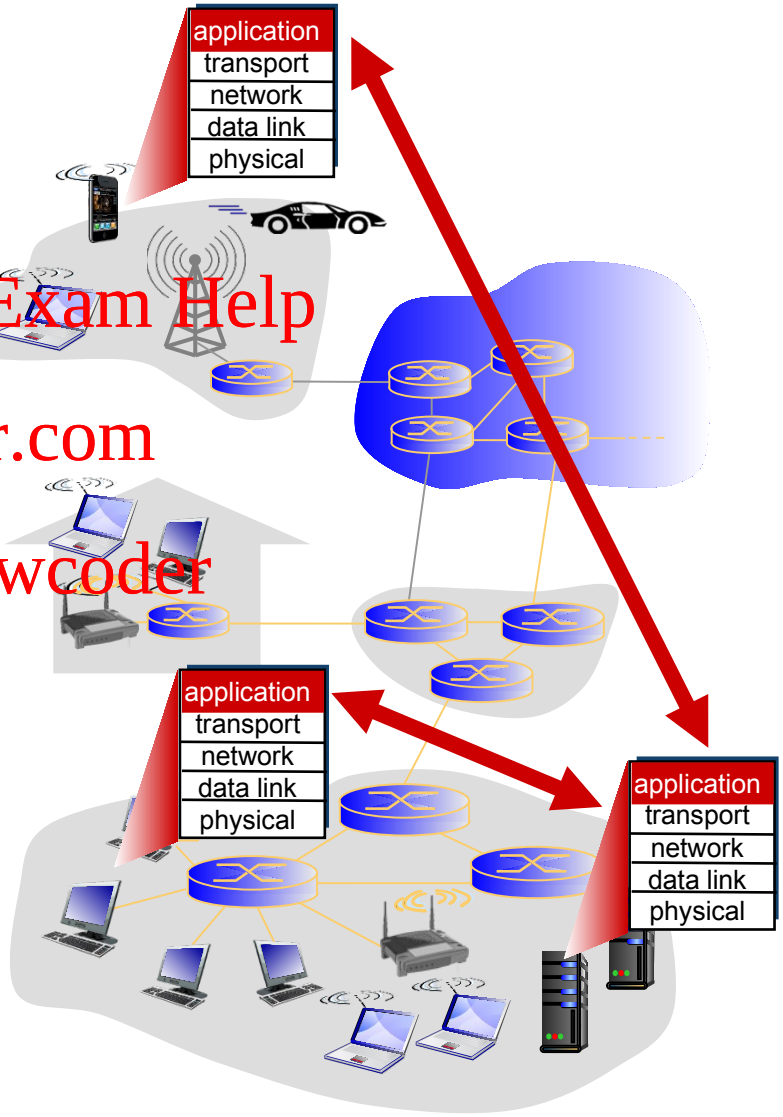
Creating a network app

write programs that:

- run on (different) *end systems*
- communicate over network
- e.g., web server software communicates with browser software

no need to write software for network-core devices

- network-core devices do not run user applications
- applications on end systems allows for rapid app development, propagation

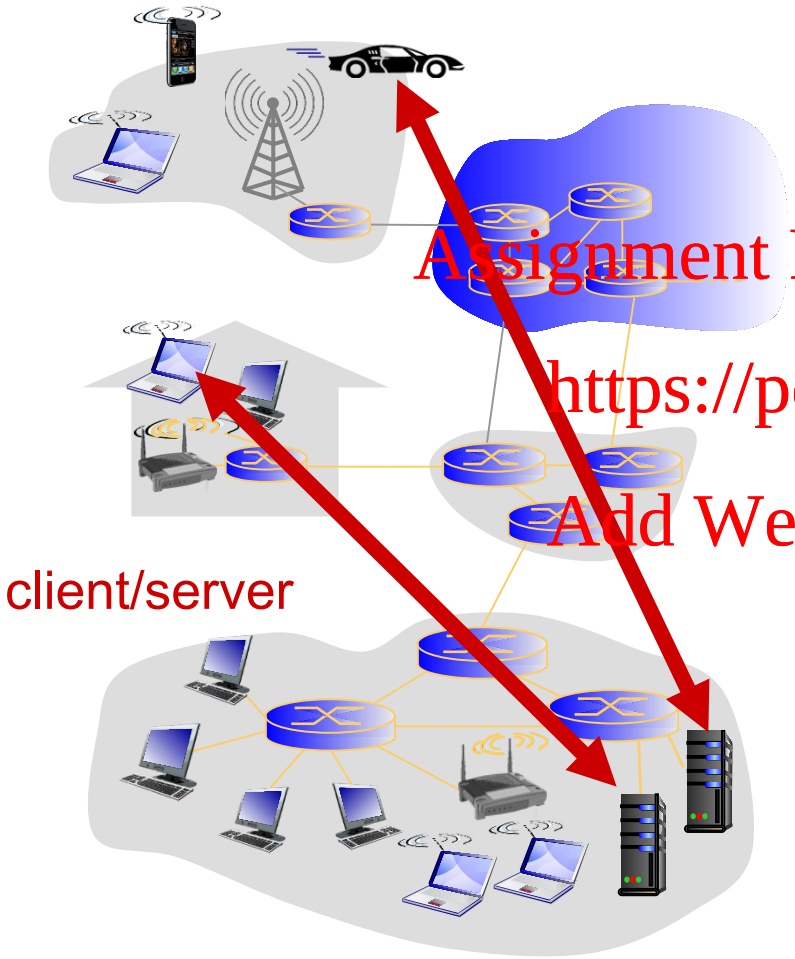


Application architectures

Possible structure of applications

- Client-server
- Peer-to-peer (P2P)

Client-server architecture



server:

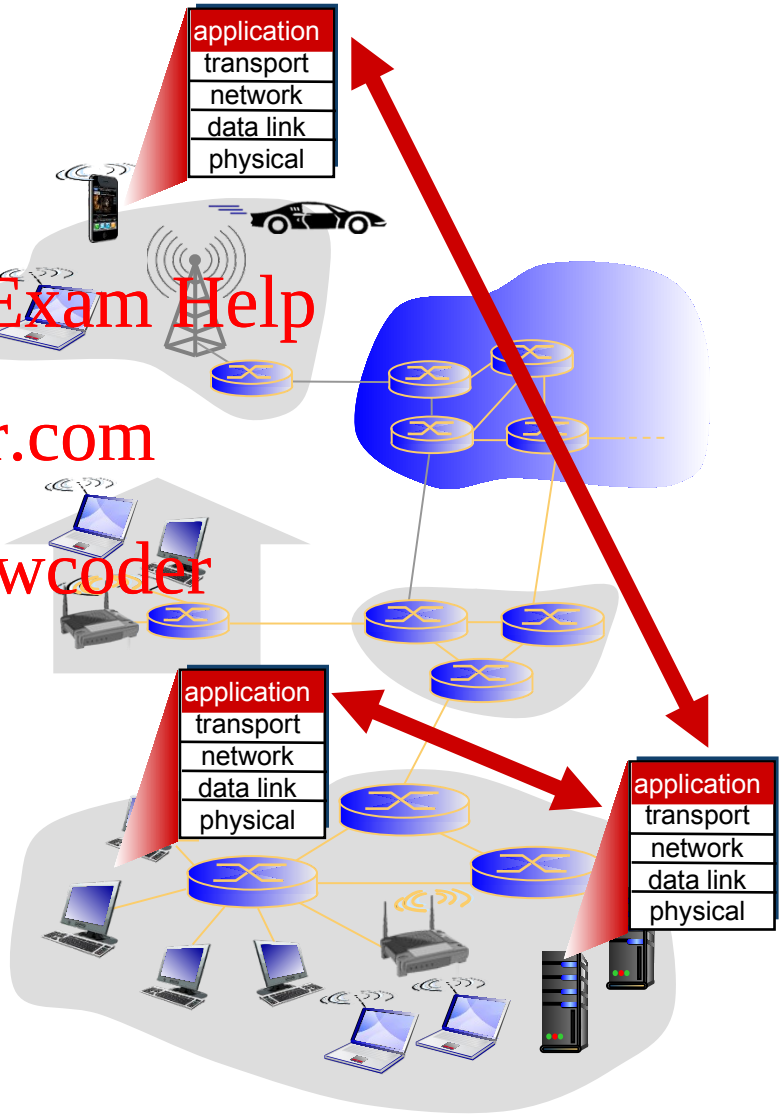
- always-on
- permanent IP address
- data centers for scaling

clients:

- communicate with server
- may be intermittently connected
- may have dynamic IP addresses
- do not communicate directly with each other

P2P architecture

- *no* always-on server
- arbitrary end systems directly communicate
- peers request service from other peers, provide service in return to other peers
  - *self scalability* – new peers bring new service capacity, as well as new service demands
- peers are intermittently connected and change IP addresses
  - complex management



Process communicating

process: program running within a host

- within same host, two processes communicate using inter-process communication (defined by OS)
- processes in different hosts communicate by exchanging messages

clients, servers

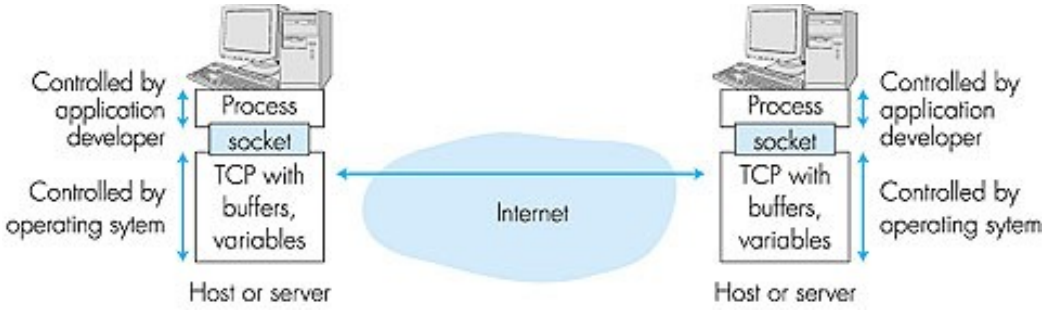
client process: process that initiates communication

server process: process that waits to be contacted

- aside: applications with P2P architectures have client processes & server processes

Sockets

- process sends/receives messages to/from its socket
- socket analogous to door
  - sending process shoves message out door
  - sending process relies on transport infrastructure on other side of door to deliver message to socket at receiving process



Addressing processes

- to receive messages, process must have *identifier*
- host device has unique 32-bit IP address (or 128 in IPv6)
- *Q:* does IP address of host on which process runs suffice for identifying the process?
  - *A:* no, *many* processes can be running on same host

- *identifier* includes both IP address and port numbers associated with process on host.
- example port numbers:
  - HTTP server: 80
  - mail server: 25
- to send HTTP message to gaia.cs.umass.edu web server:
  - IP address: 128.119.245.12
  - port number: 80
- more shortly…

App-layer protocol defines

- types of messages exchanged,
  - e.g., request, response
- message syntax:
  - what fields in messages & how fields are delineated
  - e.g. First line: method. Second line: URL
  - message semantics
  - meaning of information in fields
  - e.g. 404 means "not found"
- rules for when and how processes send & respond to messages

open protocols:

- defined in RFCs
- allows for interoperability
- e.g., HTTP, SMTP

proprietary protocols:

- e.g., Skype

What transport service does an app need?

data integrity

- some apps (e.g., file transfer, web transactions) require 100% reliable data transfer
- other apps (e.g., audio) can tolerate some loss

timing

- some apps (e.g., Internet telephony, interactive games) require low delay to be "effective"

throughput

- some apps (e.g., multimedia) require minimum amount of throughput to be "effective"
- other apps ("elastic apps") make use of whatever throughput they get

Internet transport protocols services

TCP service:

- *reliable transport* between sending and receiving process
- *flow control:* sender won't overwhelm receiver
- *congestion control:* throttle sender when network overloaded
- *does not provide:* timing, minimum throughput guarantee
- *connection-oriented:* setup required between client and server processes

UDP service:

- *unreliable data transfer* between sending and receiving process
- *does not provide:* reliability, flow control, congestion control, timing, throughput guarantee, or connection setup,

Internet apps: application, transport protocols

application	application layer protocol	underlying transport protocol
email	SMTP [RFC 2821]	TCP
remote terminal access	Telnet [RFC 854]	TCP
Web	HTTP [RFC 2616]	TCP
file transfer	FTP [RFC 959]	TCP
streaming multimedia	HTTP RTP [RFC 1889]	TCP or UDP
Internet telephony	SIP, RTP, proprietary (e.g., Skype)	TCP or UDP

Web and HTTP

First, a review…

- web page consists of *base HTML-file* which includes *several referenced objects*
  - *HTML: HyperText Markup Language*
- object can be JPEG image, Java applet, audio file,…
- each object is addressable by a *URL (Uniform Resource Locator),* e.g.,

`www.someschool.edu/someDept/pic.gif`

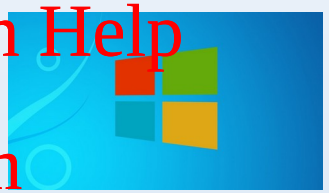
host name | path name

File: usually base-html file
(HyperText Markup Language)

xxxxxxxxx
www.aaa.edu/Obj1.jpg
yyyyyyyyyyy
www.aaa.edu/Obj2.jpg
zzzzzzzzz

Browser shows

xxxxxxxxx



yyyyyyyyyyy

zzzzzzzzz

HTTP: hypertext transfer protocol

- Web's application layer protocol
- client/server model
  - *client:* browser that requests, receives, (using HTTP protocol) and "displays" Web objects
  - *server:* Web server sends (using HTTP protocol) objects in response to requests

HTTP overview (cont'd)

uses TCP:

- client initiates TCP connection (creates socket) to server, port 80
  - How to know IP address?
  - DNS (Domain Name System)
- server accepts TCP connection from client
- HTTP messages (application-layer protocol messages) exchanged between browser (HTTP client) and Web server (HTTP server)
- TCP connection closed

HTTP is "stateless"

- server maintains no information about past client requests

aside

protocols that maintain "state" are complex!

- past history (state) must be maintained
- if server/client crashes, their views of "state" may be inconsistent, must be reconciled

HTTP connections

non-persistent HTTP

- at most one object sent over TCP connection
  - connection then closed
- downloading multiple objects required multiple connections

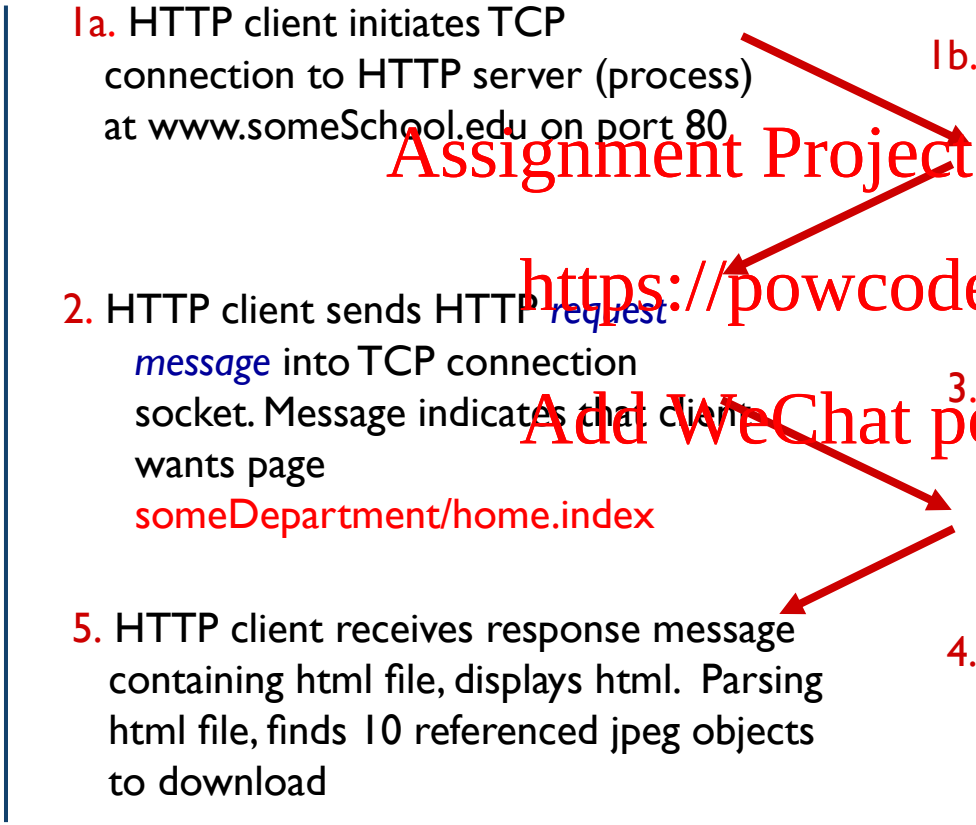
persistent HTTP

- multiple objects can be sent over single TCP connection between client, server

Non-persistent HTTP

suppose user enters URL:
`www.someSchool.edu/someDepartment/home.index`

(contains text, references to 10 jpeg images)



Non-persistent HTTP

suppose user enters URL:
`www.someSchool.edu/someDepartment/home.index`

(contains text, references to 10 jpeg images)

1a. HTTP client initiates TCP connection to HTTP server (process) at www.someSchool.edu on port 80

1b. HTTP server at host www.someSchool.edu waiting for TCP connection at port 80. "accepts" connection, notifying client

2. HTTP client sends HTTP *request message* into TCP connection socket. Message indicates that client wants object someDepartment/object1.jpg

3. HTTP server receives request message, forms *response message* containing requested object, and sends message

5. HTTP client receives response message containing object, displays the object.

4. HTTP server closes TCP connection.

time

6. Steps 1-5 repeated for each of 10 jpeg objects

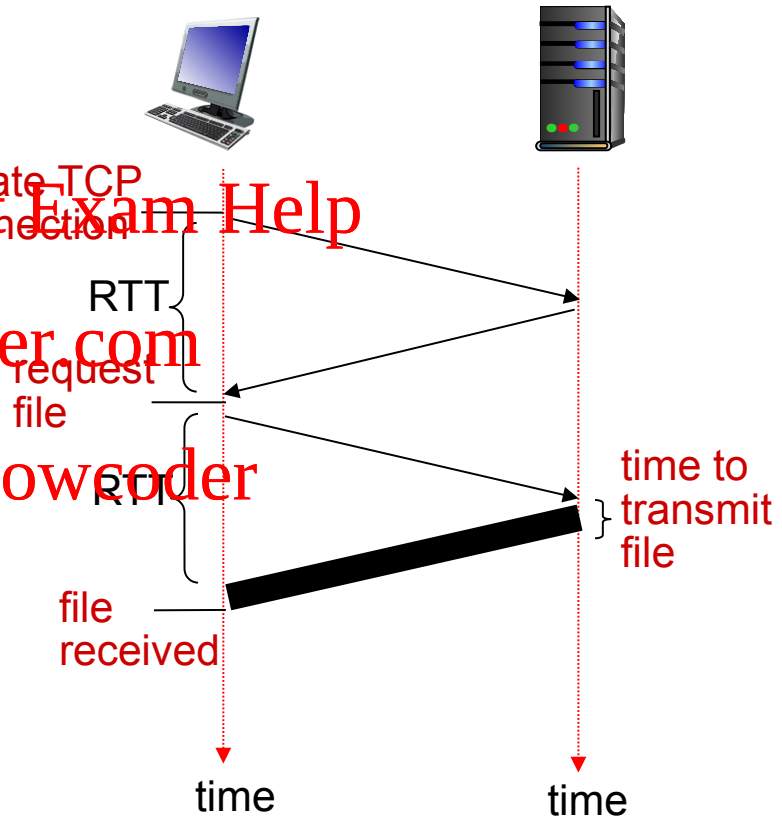
HTTP: response time

RTT (definition): time for a small packet to travel from client to server and back

HTTP response time:

- one RTT to initiate TCP connection
- one RTT for HTTP request and first few bytes of HTTP response to return
- file transmission time
- non-persistent HTTP response time =

  2RTT+ file transmission time



Persistent HTTP

suppose user enters URL:

`www.someSchool.edu/someDepartment/home.index`

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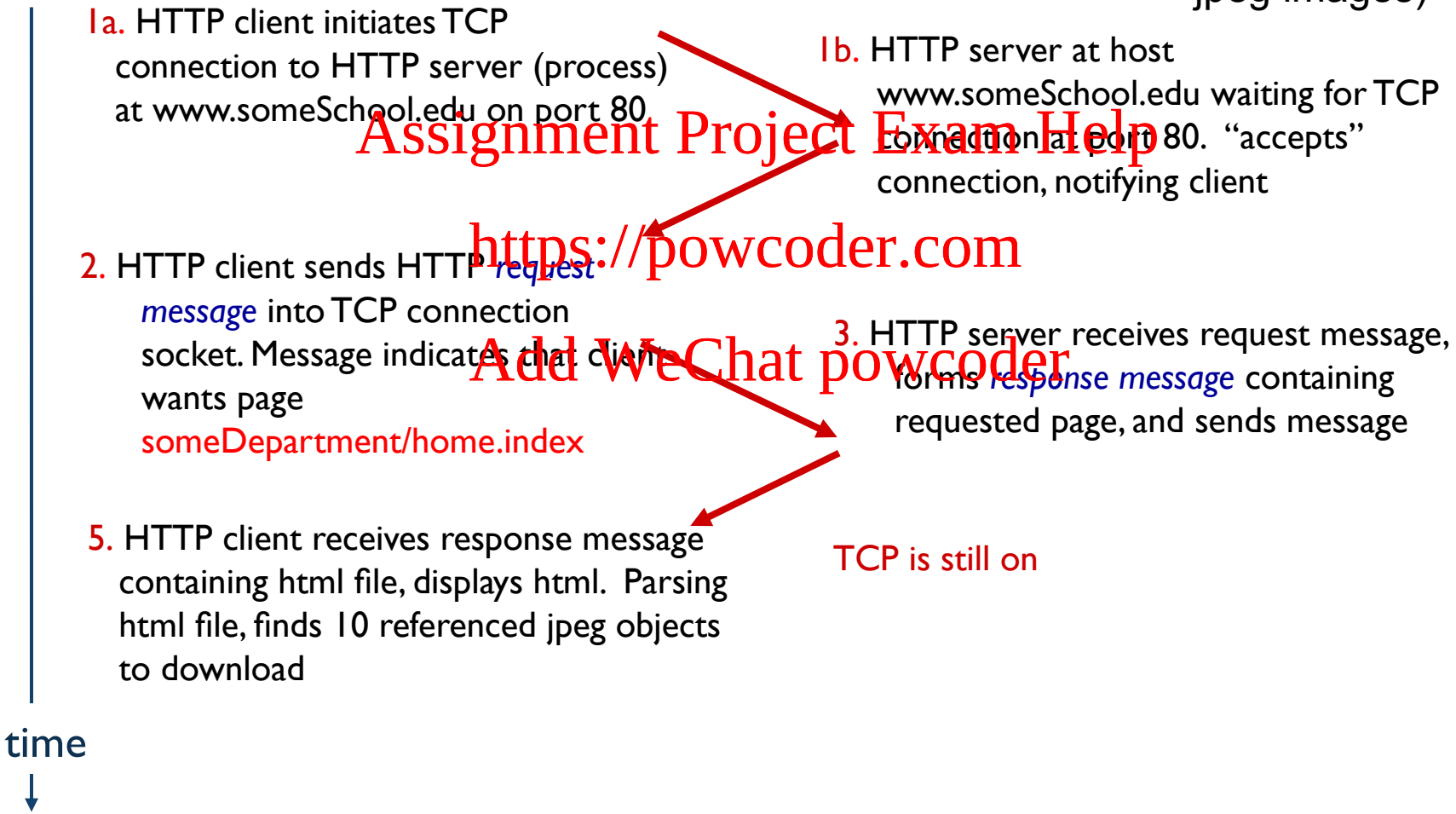


Persistent HTTP

suppose user enters URL:
`www.someSchool.edu/someDepartment/home.index`

(contains text, references to 10 jpeg images)

2. HTTP client sends HTTP *request message* into TCP connection socket. Message indicates that client wants object someDepartment/object1.jpg

3. HTTP server receives request message, forms *response message* containing requested object, and sends message

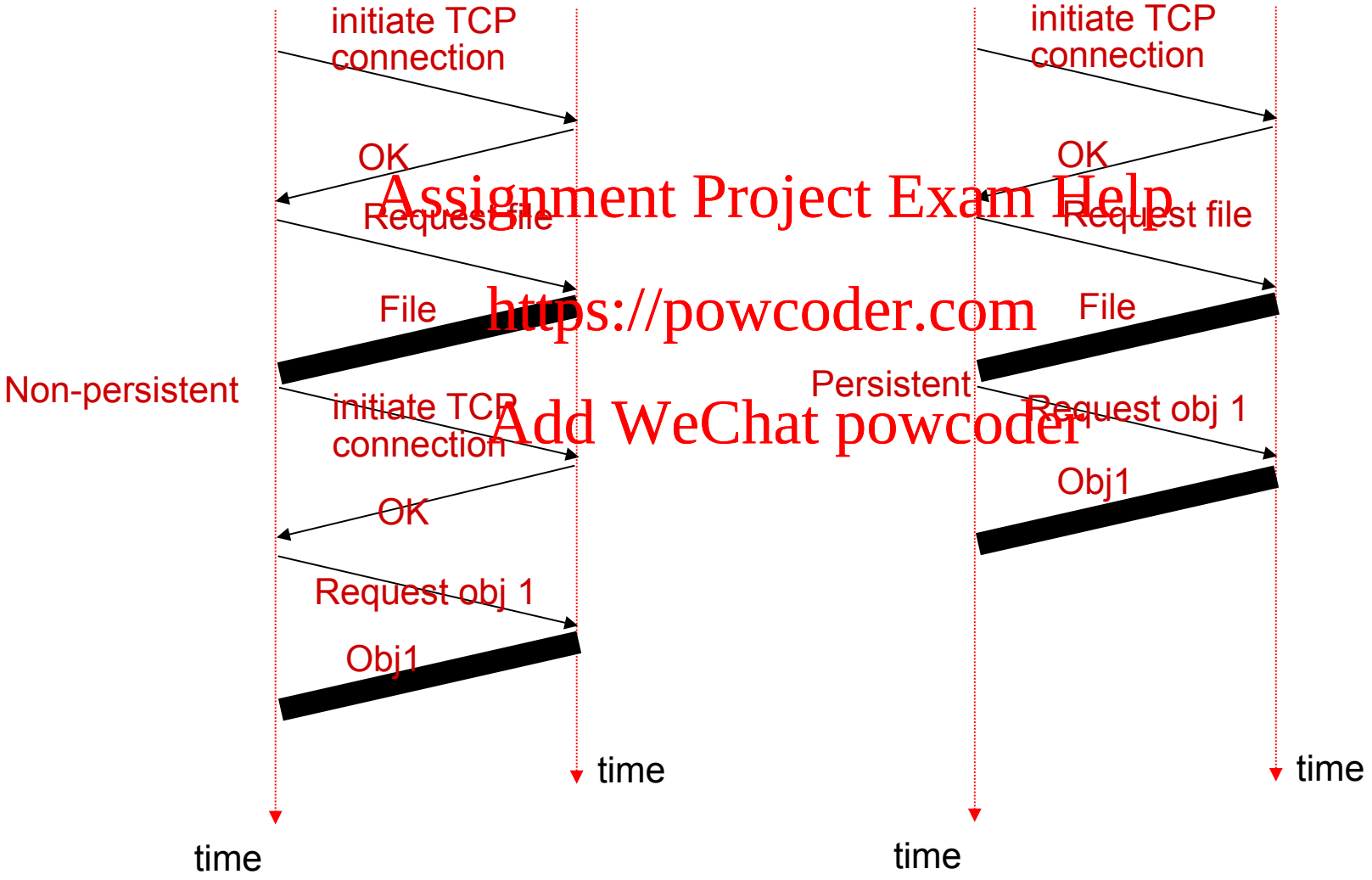
4. HTTP client receives response message containing object, displays the object.

Repeated for each of 10 jpeg objects

10 rounds later HTTP server closes TCP connection.

time

Non-persistent vs. persistent



Persistent HTTP

non-persistent HTTP issues:

- requires 2 RTTs + file transmission time per object

persistent HTTP:

- server leaves connection open after sending response
- subsequent HTTP messages between same client/server sent over open connection
- client sends requests as soon as it encounters a referenced object
- as little as one RTT + file transmission time for all the referenced objects

HTTP request message

- two types of HTTP messages: *request, response*
- HTTP request message:
  - ASCII (human-readable format)

carriage return character

line-feed character

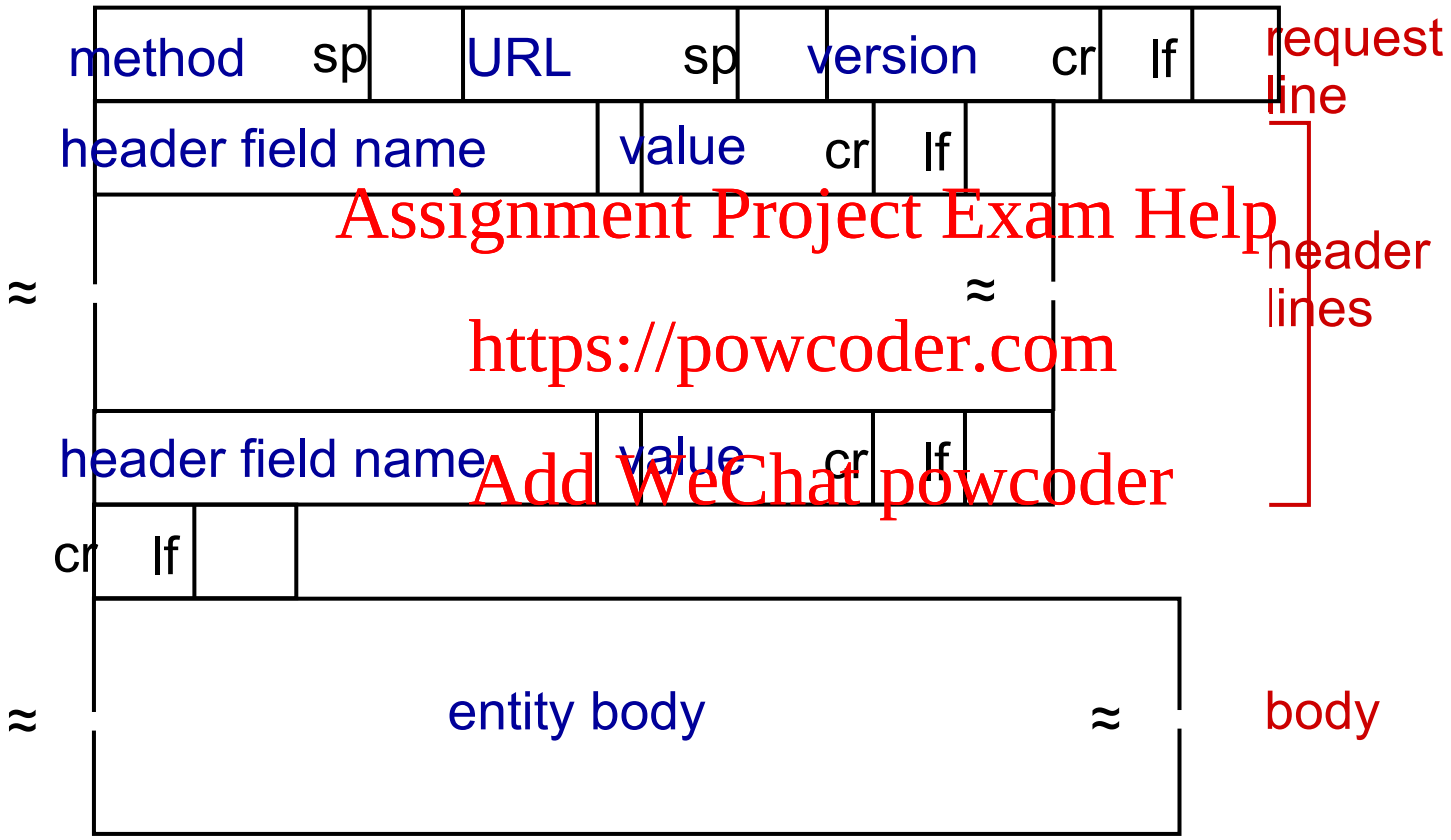
request line (GET, POST, HEAD commands)


```
GET /index.html HTTP/1.1\r\n
Host: www-net.cs.umass.edu\r\n
User-Agent: Firefox/3.6.10\r\n
Accept: text/html,application/xhtml+xml\r\n
Accept-Language: en-us,en;q=0.5\r\n
Accept-Encoding: gzip,deflate\r\n
Accept-Charset: ISO-8859-1,utf-8;q=0.7\r\n
Keep-Alive: 115\r\n
Connection: keep-alive\r\n
\r\n
```

header lines

carriage return, line feed at start of line indicates end of header lines

HTTP request message: general format



Uploading form input

GET method

POST method:

- web page often includes form input
- input is uploaded to server in entity body

Method types

HTTP/1.0:

- GET
- POST
- HEAD
  - asks server to leave requested object out of response

HTTP/1.1:

- GET, POST, HEAD
- PUT
  - uploads file in entity body to path specified in URL field
- DELETE
  - deletes file specified in the URL field

HTTP response message

status line (protocol status code status phrase)

header lines

data, e.g., requested HTML file

```
HTTP/1.1 200 OK\r\n
Date: Sun, 26 Sep 2010 20:09:20 GMT\r\n
Server: Apache/2.0.52 (CentOS)\r\n
Last-Modified: Tue, 30 Oct 2007 17:00:02
   GMT\r\n
ETag: "17dc6-a5c-bf716880"\r\n
Accept-Ranges: bytes\r\n
Content-Length: 2652\r\n
Keep-Alive: timeout=10, max=100\r\n
Connection: Keep-Alive\r\n
Content-Type: text/html; charset=ISO-8859-1\
   r\n
\r\n
data data data data data ...
```

HTTP response status codes

- status code appears in 1st line in server-to-client response message.
- some sample codes:

200 OK

- request succeeded, requested object later in this msg

301 Moved Permanently

- requested object moved, new location specified later in this msg (Location:)

400 Bad Request

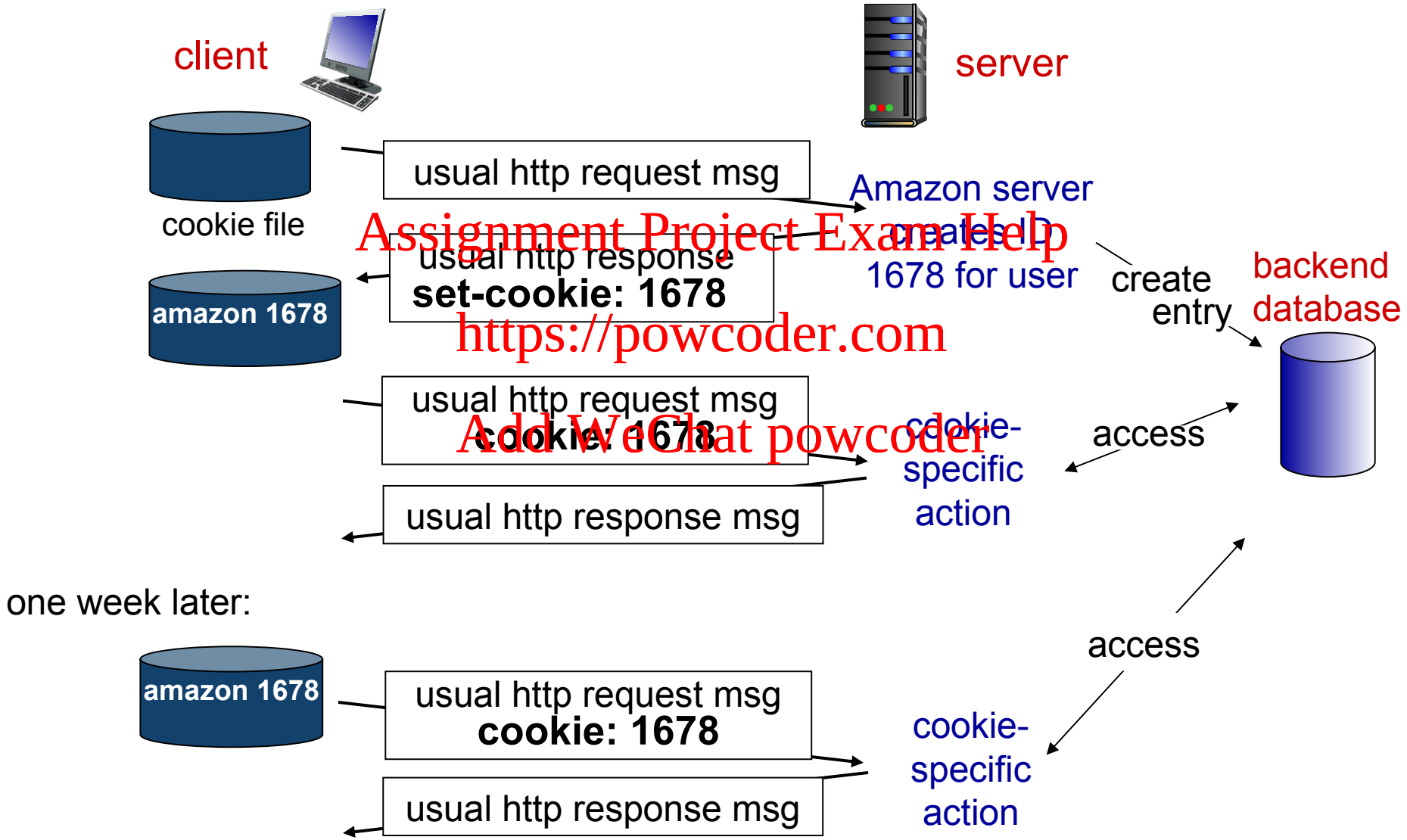
- request msg not understood by server

404 Not Found

- requested document not found on this server

505 HTTP Version Not Supported

Cookies: keeping “state” (cont’d)



User-server state: cookies

many Web sites use cookies

four components:

1) cookie header line of HTTP *response* message
2) cookie header line in next HTTP *request* message
3) cookie file kept on user's host, managed by user's browser
4) back-end database at Web site

Cookies (cont'd)

what cookies can be used for:

› authorization

› shopping carts

› recommendations

› user session state (Web e-mail)

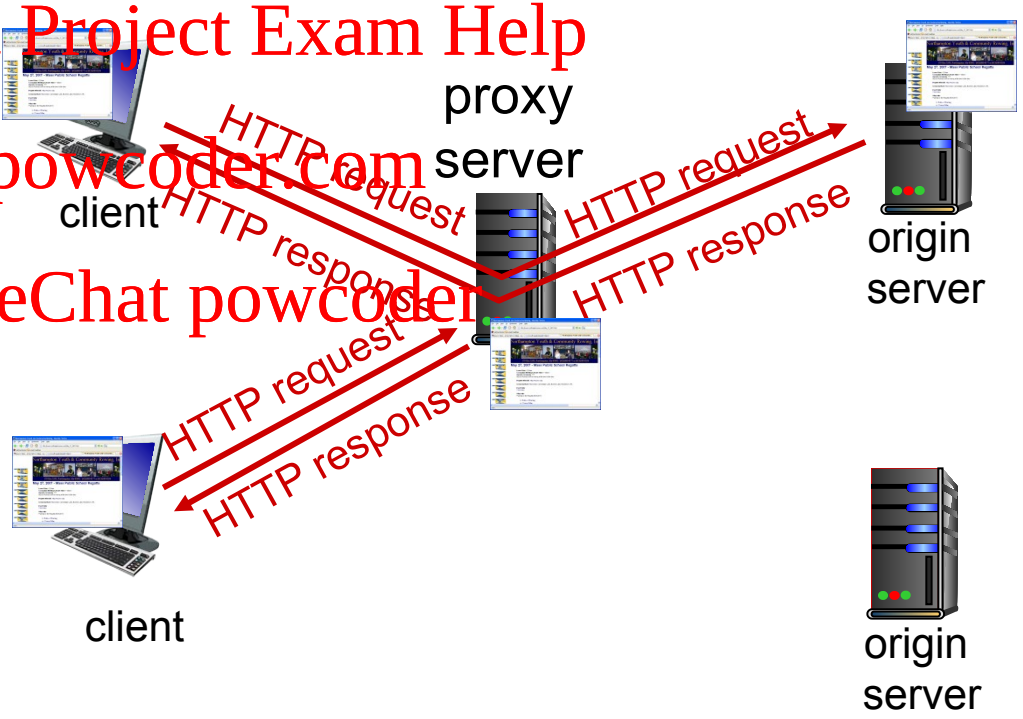
how to keep "state":

- protocol endpoints: maintain state at sender/receiver over multiple transactions
- cookies: http messages carry state

Web caches (proxy server)

goal: satisfy client request without involving origin server

- user sets browser: Web accesses via cache
- browser sends all HTTP requests to cache
- if object in cache:
  - then cache returns object
  - else cache requests object from origin server, then returns object to client



More about Web caching

› Q: Does the cache act as a client or a server?

More about Web caching

- R: cache acts as both client and server
  - server for original requesting client
  - client to origin server
- typically cache is installed by ISP (university, company, residential ISP)

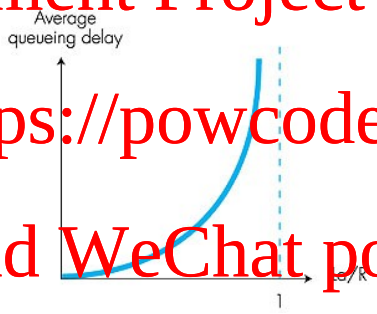
why Web caching?

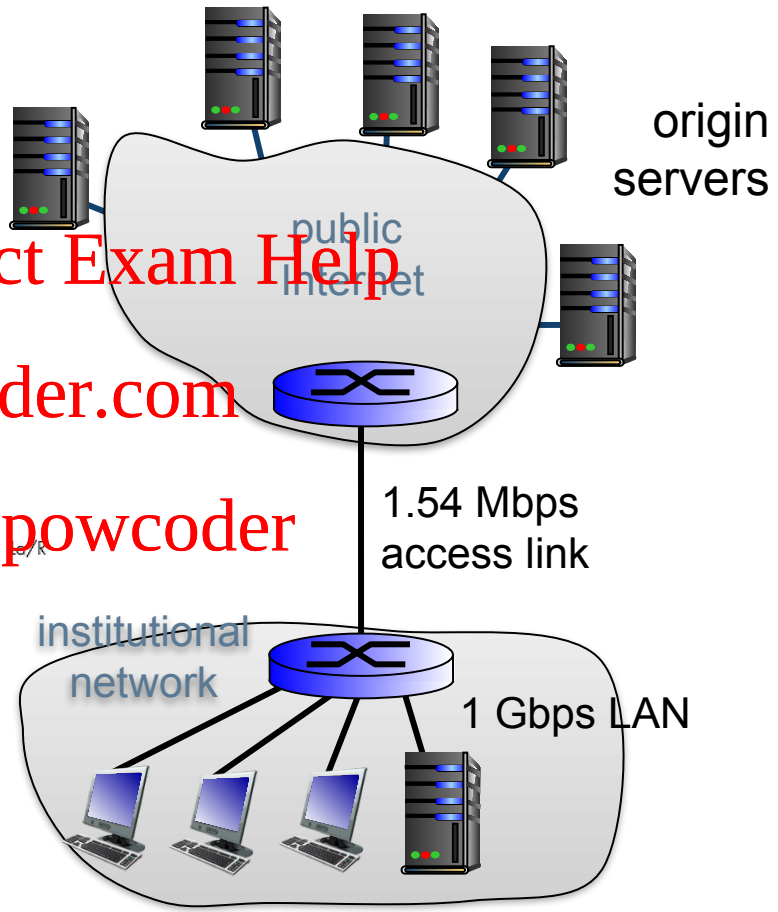
- reduce response time for client request
- reduce traffic on an institution's access link

Caching example

assumptions:

- avg object size: 100K bits
- avg request rate from browsers to origin servers: 15/sec (1.5 Mbps service)
- RTT from institutional router to any origin server: 2 sec
- access link rate: 1.54 Mbps





consequences:

- LAN utilization: 0.15%
- LANU = avg req rate * size / link bandwidth
- access link utilization = 99% *problem!*
- ALU = avg req rate * size / link bankwidth
- total delay = 2 sec + minutes + usecs

Q: what happens with fatter access link?

Caching example: fatter access link

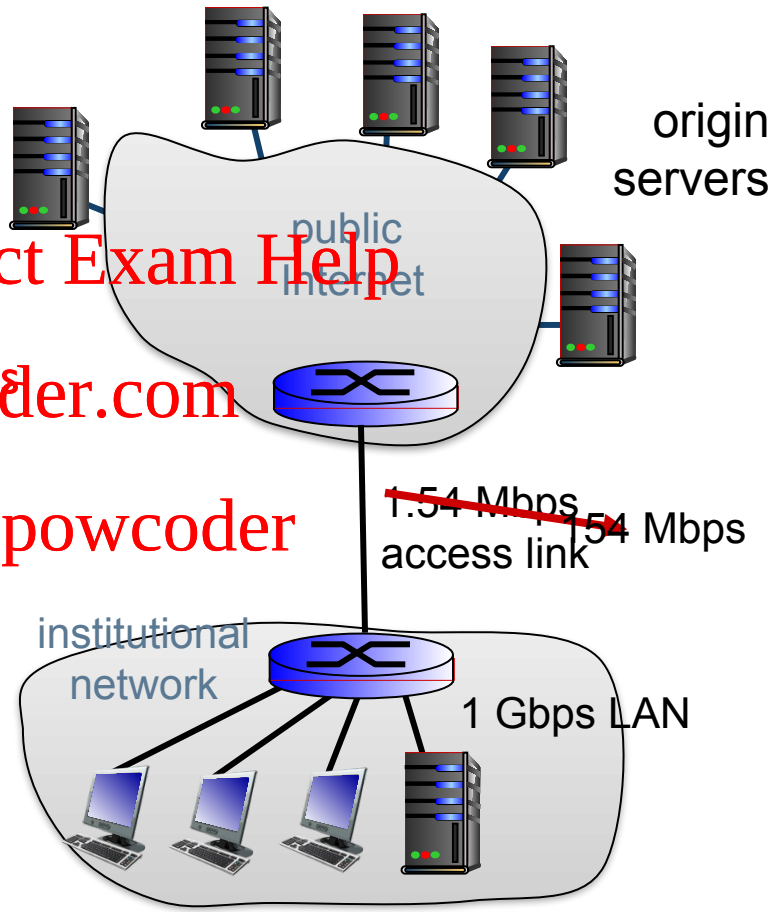
assumptions:

- avg object size: 100K bits
- avg request rate from browsers to origin servers: 15/sec
- RTT from institutional router to any origin server: 2 sec
- access link rate: 1.54 Mbps

consequences:

- LAN utilization: 0.15%
- access link utilization = 99%
- total delay = 2 sec + minutes + usecs

msecs



Cost: increased access link speed (not cheap!)

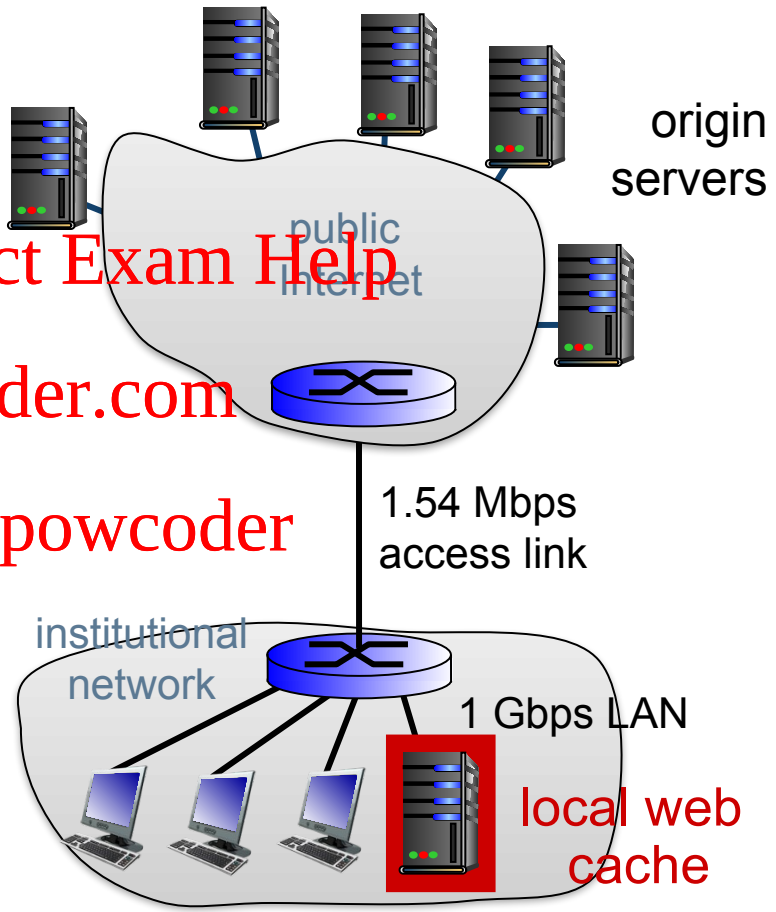
Caching example: install local cache

assumptions:

- avg object size: 100K bits
- avg request rate from browsers to origin servers: 15/sec
- RTT from institutional router to any origin server: 2 sec
- access link rate: 1.54 Mbps

consequences:

- LAN utilization: 0.15%
- access link utilization = 0%
- total delay = usecs



Cost: web cache (cheap!)

Caching example: install local cache

Calculating access link utilization, delay with cache:

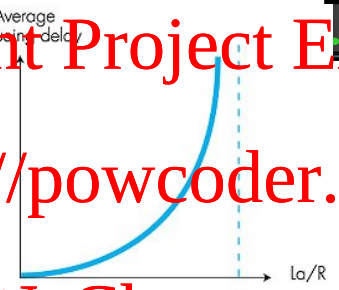
- suppose cache hit rate is 0.4
  - 40% requests satisfied at cache
  - 60% requests satisfied at origin
- access link utilization:
  - 60% of requests use access link
- average total delay

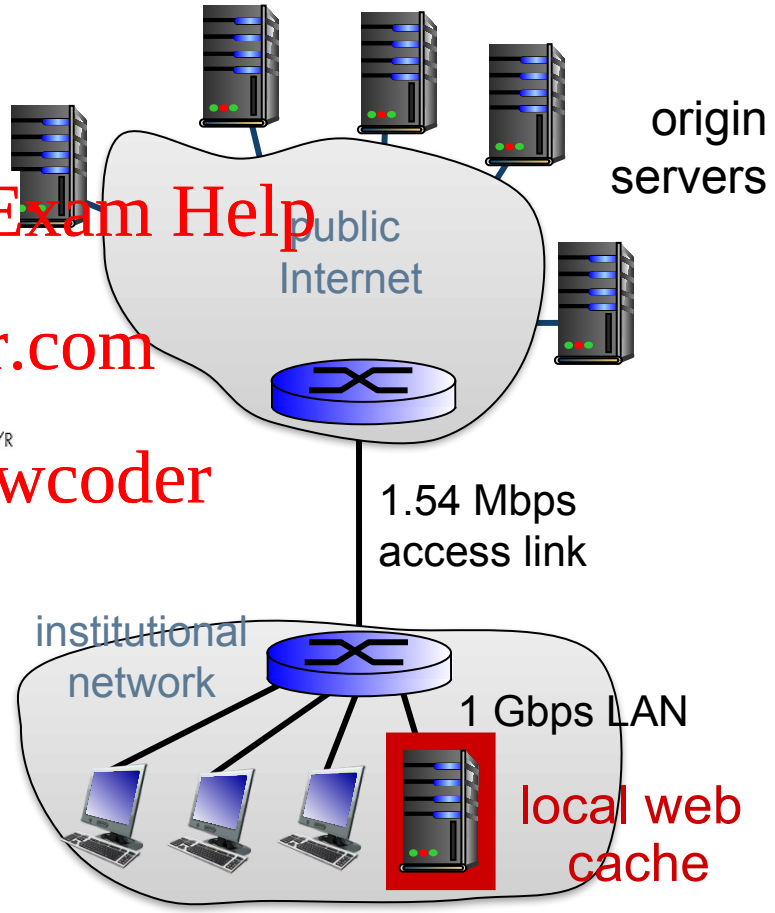
  = 0.6 * (delay from origin servers) +0.4 * (delay when satisfied at cache)

  Link utilization is around 60%, queueing delay is small enough

  = 0.6 (~2.x second) + 0.4 (~usecs)

  less than with 154 Mbps link (and cheaper too!)





Conditional GET

- *Goal:* don't send object if client has up-to-date cached version
  - no object transmission delay
  - lower link utilization
- *client:* specify date of cached copy in HTTP request

  `If-modified-since: <date>`
- *server:* response contains no object if cached copy is up-to-date:

  `HTTP/1.0 304 Not Modified`